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2016.07 Cisco Official News 300-075 CIPTV2:Implementing Cisco IP Telephony & Video, Part 2 v1.0 Exam Questions 355q Updated by Braindump2go.com for Free Instant Download Today! 100% 300-075 Exam Pass Guaranteed! NEW QUESTION 61 -NEW OUESTION 70: 2016 100% Valid Cisco 300-075 PDF & 300-075 VCE Dumps

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ip host \_cisco-uds.\_tcp.hq.cisco.com srv 1 1 8443 10.1.5.16 Brillia 216 com ip host publox-hq.hq.collablox.cisco.com 10.1.5.15 ip host sublox-hq.hq.collablox.cisco.com 10.1.5.16 ip host hq.cisco.com 10.1.5.1

Device Pool:



Expressway:





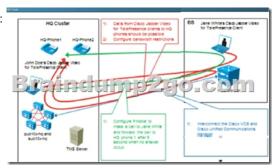


MRA: ump2go.com

Speed Dial:

Phone	Speed Dial Button	Speed Dial Destination
HQ phone 1	1	hq2@cisco.lab
HQ phone 1	2	br1@cisco.lab
HQ phone 1	3	bb@cisco.lab
HQ phone 2	1	hq1@cisco.lab
Pphone 2 Inclum	ip2g	br1@cisco.lab br1@cisco.lab
HQ phone 2	3	bb@cisco.lab
BR1 phone 1	1	hq1@cisco.lab
BR1 phone 1	2	hq2@cisco.lab
BR1 phone 1	3	bb@cisco.lab
BB phone	1	hq1@cisco.lab
BB phone	2	hq2@cisco.lab

## SIP TRUNK:



Which three configuration tasks need to be completed on the router to support the registration of Cisco Jabber clients? (Choose three.) A. The DNS server has the wrong IP address.B. The internal DNS Service (SRV) records need to be updated on the DNS Server.C. Flush the DNS Cache on the client.D. The DNS AOR records are wrong.E. Add the appropriate DNS SRV for the Internet entries on the DNS Server. Answer: BCE QUESTION 63With Media Gateway Control Protocol configuration on the voice gateway, which three types of messages are involved in the call flow between the call agent and the voice gateway? (Choose three.) A. audit endpointB. modify endpointC. create connectionD. delete notificationE. restart in progressF. end connections Answer: ACE QUESTION 64Which task must you perform before deleting a transcoder? A. Delete the dependency records.B. Unassign it from a media resource group.C. Use the Reset option.D. Remove the device pool.E. Remove the subunit.F. Delete the common device configuration. Answer: B QUESTION 65How many active gatekeepers can you can define in a local zone? A. 1B. 2C. 5D. 10E. 15F. unlimited Answer: A QUESTION 66Company X has a Cisco Unified Communications Manager cluster and a VCS Control server with video endpoints registered on both systems. Users find that video endpoints registered on Call manager can call each other and likewise for the endpoints registered on the VCS server. The administrator for Company X realizes he needs a SIP trunk between the two systems for any video endpoint to call any other video endpoint. Which two steps must the administrator take to add the SIP trunk? (Choose two.) A. Set up a SIP trunk on Cisco UCM with the option Device-Trunk with destination address of the VCS server.B. Set up a subzone on Cisco UCM with the peer address to the VCS cluster.C. Set up a neighbor zone on the VCS server with the location of Cisco UCM using the menu option VCS Configuration > Zones > zone.D. Set up a SIP trunk on the VCS server with the destination address of the Cisco UCM and Transport set to TCP.E. Set up a traversal subzone on the VCS server to allow endpoints that are registered on Cisco UCM to communicate. Answer: AC QUESTION 67Scenario: There are two call control systems in this item. The Cisco UCM is controlling the DX650. the Cisco Jabber for Windows Client, and the 9971 Video IP Phone. The Cisco VCS and TMS controll the the Cisco TelePresence MCU, and the Cisco Jabber TelePresence for WindowsDP:

Locations:







#### SRST:



## SRST-BR2-Config:

- Name: BR2
- Port: 2000
- IP Address: 10.1.5.15
- SIP Network/IP Address: 10.1.5.15

## BR2 Config:

```
voice service voip
  bind control source-interface
GigabitEthernet0/0/0.130
  bind media source-interface
GigabitEthernet0/0/0.130
registrar server
Dice register global
 max-pool 1
voice register pool 1
 id network 10.1.130.0 mask 255.255.25
call-manager-fallback
 ip source-address 10.1.130.1
 max-dn 1 dual-line
 max-ephones 1
```

```
SESTPSTNCall: At the HQ cluster, the CFUR for the directory number that is applied to BR2 phone (+442288224001) has been configured:
```

- Forward Unregistered Internal Destination: +442288224001 ■ Colward Unregisted Livershell Cal Colored Space: System & M
- Forward Unregistered External Destination: +442288224001
- Forward Unregistered External Calling Search Space: System css

After configuring the CFUR for the directory number that is applied to BR2 phone (+442288224001). the calls fail from the PSTN. Which two of the following configurations if applied to the router, would remedy this situation? (Choose two.) A. dial-peer voice 1 potsincoming called-number 228822....direct-inward-dialport 0/0/0:15B. dial-peer voice 1 potsincoming called-number 228822.... direct-inward-dialport 0/0/0:13C. voice translation-rule 1rule 1/rt228821....S//+44&/exit! voice translation-profile pstn-intranslate called 1!voice-port 0/0/0:15translation-profile incoming pstn-inD. voice translation-rule 1rule 1/rt228822....S//+44&/exit! voice translation-profile pstn-intranslate called 1!voice-port 0/0/0:15translation-profile incoming pstn-inE. The router does not need to be configured. Answer: AD QUESTION 68Refer to the exhibit. Which configuration change is needed to enable NANP international dialing during MGCP fallback? A. Change the dial peer to dial-peer voice 901 voip.B. Change the dial peer to dial-peer voice 9011 pots.C. Add the command prefix 011 to the dial peer.D. Add the command prefix 9011 to the dial peer. Answer: C QUESTION 69Which statement is true regarding the configuration of SAF Forwarder? A. In a multisite dial plan, SAF Forwarders may exist in multiple autonomous systems.B. The client label that is configured in Cisco Unified Communications Manager must match the configuration on the SAF Forwarder router.C. There should not be multiple nodes of Cisco Unified Communications Manager clusters acting as SAF clients.D. The destination IP address must match the loopback address of the SAF router. Answer: A QUESTION 70Which technologies provide remote-site redundancy for Cisco IP Phones during a WAN failure? A. SRST and MGCP fallbackB. SRST and TEHOC. TEHO and MGCP fallbackD. SRST and AAR Answer: A !!!RECOMMEND!!! 1.]Braindump2go 2016.07 New Cisco 300-075 Exam PDF and VCE 355Q&As Dumps Instant Download:http://www.braindump2go.com/300-075.html [100% Exam Pass Promised!] 2.|Braindump2go 2016.07 New Cisco 300-075 Exam Questions PDF:https://drive.google.com/folderview?id=0BwwEErkugSaLZE0xZjQtcVgyNTA&usp=sharing